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UTILITY PATENT APPLICATION TRANSMITTAL

(Only for new nonprovisional applications under 37 CFR 1.53(b))

Attorney Docket No. 081862.P164

Total Pages 5

First Named Inventor or Application Identifier Prasad Miriyala

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ADDRESS TO: Assistant Commissioner for Patents
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APPLICATION ELEMENTS

See MPEP chapter 600 concerning utility patent application contents.

1. x Fee Transmittal Form
(Submit an original, and a duplicate for fee processing)
2. x Specification (Total Pages 16)
(preferred arrangement set forth below)
 - Descriptive Title of the Invention
 - Cross References to Related Applications
 - Statement Regarding Fed sponsored R & D
 - Reference to Microfiche Appendix
 - Background of the Invention
 - Brief Summary of the Invention
 - Brief Description of the Drawings (if filed)
 - Detailed Description
 - Claims
 - Abstract of the Disclosure
3. x Drawings(s) (35 USC 113) (Total Sheets 2)
4. x Oath or Declaration (Total Pages 4)
 - a. x Newly Executed (Original or Copy)
 - b. Copy from a Prior Application (37 CFR 1.63(d))
(for Continuation/Divisional with Box 17 completed) (**Note Box 5 below**)
 - i. DELETIONS OF INVENTOR(S) Signed statement attached deleting inventor(s) named in the prior application, see 37 CFR 1.63(d)(2) and 1.33(b).
5. Incorporation By Reference (useable if Box 4b is checked)
The entire disclosure of the prior application, from which a copy of the oath or declaration is supplied under Box 4b, is considered as being part of the disclosure of the accompanying application and is hereby incorporated by reference therein.
6. Microfiche Computer Program (Appendix)

[illegible]

8. x Assignment Papers (cover sheet & documents(s))

9. a. 37 CFR 3.73(b) Statement (where there is an assignee)

 x b. Power of Attorney

10. English Translation Document (if applicable)

11. — a. Information Disclosure Statement (IDS)/PTO-1449

 b. Copies of IDS Citations

12. Preliminary Amendment

13. x Return Receipt Postcard (MPEP 503) (Should be specifically itemized)

14. a. Small Entity Statement(s)

 b. Statement filed in prior application, Status still proper and desired

15. Certified Copy of Priority Document(s) (if foreign priority is claimed)

16. x Other: Copy of Postcard w/Express Mail Stamp

____ Continuation ____ Divisional ____ Continuation-in-part (CIP)
of prior application No: ____

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UNITED STATES PATENT APPLICATION

for

**METHOD FOR SUPPORTING HIGH PRIORITY CALLS ON A CONGESTED
WAN LINK**

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METHOD FOR SUPPORTING HIGH PRIORITY CALLS ON A CONGESTED WAN LINK

Field of the Invention

5 This invention relates to resource allocation within a computer or other network. More particularly, the present invention relates to the allocation of bandwidth in a Digital Communications Network.

Background

10 **Figure 1** generally depicts a digital network **100** adapted to transport voice, data and other information. This exemplary network will be used to illustrate some of the drawbacks associated with present networking communication methodologies and equipment. The network includes a plurality of nodes **102, 104** (hereinafter “network nodes”), each of which may be coupled to various customer premises equipment (CPE) (not shown) and/or other network nodes (not shown). Network nodes **102** and **104** are
15 communicatively coupled to one another through a communication link **106**, which may be a wired or wireless communication link. In this network **100**, nodes **102** and **104** communicate across communication link **106** using the well-known Asynchronous Transfer Mode (ATM) protocol. Thus, at each end of communication link **106**, ATM
20 interfaces **108** are employed. Communication between nodes **102** and **104** with various CPE and/or other nodes is supported across communication links **108** and **110**, respectively, each of which is associated with an appropriate interface **112, 114** at its respective node.

When a communication (e.g., a voice communication) is to be transported within network **100**, the signal first travels from the associated CPE equipment (e.g., a private branch exchange or PBX) to node **102** over communication link (e.g., digital T1 carrier) **108**. Before being transmitted over such a T1 carrier, the signal is sampled and converted to a digital signal. A common sampling rate used with voice communications is 8000 samples per second, with each digital sample represented by 8 bits of data. Thus, the data rate of the new digital signal is: $8000 \text{ samples/sec} \times 8 \text{ bits} = 64,000 \text{ bits/sec}$. This technique is known as Pulse Code Modulation (hereinafter "PCM") and is used extensively throughout the backbone of modern telephone systems.

Although no international standard has been adopted, the T1 carrier is one method of PCM used throughout North America and Japan. The T1 carrier is comprised of 24 channels of digital data multiplexed together. Digitally sampled data from each of the 24 channels are packaged into successive frames of $8 \text{ bits/channel} \times 24 \text{ channels} + \text{an additional framing bit} = 193 \text{ bits}$. Outside of North America and Japan a similar standard, known as E1, is commonly implemented. E1 operates in a manner similar to T1 except that it uses 32, 8-bit data samples (i.e., 32 channels) instead of 24.

After the signal has been sampled, converted to a digital signal and transmitted over a T1 carrier, as described above, it is then transferred to an outgoing communication link **106**. Because the transport protocol across communication link **106** is different than that used on communication link **108**, the digital data samples are packaged according to the protocol used across communication link **106** (e.g., ATM) before being transmitted to node **104**. Additionally, although PCM by itself provides for a data transfer rate of 64,000 bits/second, it is often desirable to further compress the digital PCM data in order

to save bandwidth within the network. This can be accomplished using Digital Signal Processing (hereinafter "DSP") resources associated with network node **102**. For example, if the 64,000 bits/second PCM signal is compressed by a DSP resource at a compression ratio of 16:1, the resulting digital signal will be transmitted at 4,000

bits/second. This represents a significant reduction in required bandwidth across network **100** to transmit the same underlying signal. Such compression techniques are particularly useful in networks that are heavily loaded with network traffic. Examples of compression algorithms known in the art include the International Telegraph Union (hereinafter "ITU") standards G.711, G.726, G.729-A, G.729, and G.728. Such compression resources may be associated with the ATM interfaces **108** and may operate under the control of a node controller **116** in each of the nodes.

However, there is a tradeoff between bandwidth savings over network **100** and the implementation of costly DSP resources at the nodes. In general, the higher the compression ratio required by the compression algorithm, the more DSP resources are used up processing the compression request over a given period of time. Thus, while a single DSP resource may process up to say 16 channels of data if no compression is used (i.e., in baseline PCM mode), it may be limited to 5 channels if data is compressed at 2:1, and only 2 channels if the PCM signal is compressed at 8:1. One factor behind this limitation is the limited period of time in which the DSP resource must operate compress the data within a T1 frame before it must move on to the next frame of data. Thus, the chosen compression ratio will have a significant impact on DSP resource usage.

Following compression (if used), the data samples are delivered through network **100** to node **104**, where the data may be decompressed and passed on to other CPE or

another node. The system is bi-directional to ensure 2-way communication between the nodes.

One problem with the communication scheme adopted in network 100 occurs when communication link 106 becomes congested, that is, when there is no available

5 bandwidth to support new incoming calls from CPE coupled to node **102**. Consider, for example, a situation where multiple calls being transported between nodes **102** and **104** are using all or almost all of the available bandwidth on communication link **106**. If a high priority call (e.g., a 911 or other emergency call) is now received at network node

102, either of two scenarios is possible. First, the high priority call may be rejected

10 (dropped) in the face of no available bandwidth. Second, rather than dropping the high priority call (clearly a least acceptable solution); the nodes may be configured to drop lower priority calls in order to free up bandwidth to accommodate the high priority call.

Although this solution may allow the high priority call to proceed, it is less than

satisfactory in as much as several existing calls may be dropped to support the one new

15 call. What is needed, therefore, is a more robust mechanism for handling such situations.

Summary of the Invention

. In one embodiment, a network node is configured to negotiate for connections for high priority calls (e.g., voice calls) received at the node in the face of otherwise congested outbound communication links. The negotiation is conducted in a fashion that will preserve connections for existing calls associated with the node. For example, the negotiation may be conducted so as to cause one or more of the existing calls to consume less bandwidth over the outbound communication links than was consumed at a time prior to reception of the high priority calls. Such negotiations may be initiated depending on the availability of codec resources and/or compression schemes at the node.

Another embodiment provides a method of managing a communication link between nodes of a communication network so as to ensure connection availability for one or more high priority calls over the communication link through dynamic renegotiations of call parameters for existing calls (e.g., voice calls) transported over the communication link. The communication link may preferably support communications according to the Asynchronous Transfer Mode and the dynamic renegotiations may be negotiations of compression schemes for the calls. Such dynamic renegotiations may be supported according to codec availability (e.g., as determined according to profile information) at the nodes and may be accomplished through the exchange of OAM cells between the nodes. In such a scheme, the high priority calls may be determined as such according to database information regarding called numbers.

Still another embodiment provides a network having a number of nodes connected through one or more communication links and a resource manager configured to allocate bandwidth over the communication links to high priority calls received at one or more of

the nodes without dropping existing calls within the network. The resource manager allocates bandwidth through dynamic renegotiations of existing bandwidth utilization within the network and may be a distributed resource among the nodes of the network.

Preferably, the nodes each support multiple codec resources, which compress voice

- 5 information transmitted over the communication link. The dynamic renegotiations are supported through the exchange of OAM cells between the nodes.

Brief Description of the Drawings

The present invention is illustrated by way of example and not limitation in the figures of the accompanying drawings, in which:

FIG. 1 shows generally a data network that can be configured in accordance with the present invention.

FIG. 2 shows an example of a process for renegotiating compression or other call parameters on the fly in accordance with an embodiment of the present invention.

FIG. 3 shows an example of a database for mapping incoming calls to outbound channels at a node of a communication network in accordance with an embodiment of the present invention.

Detailed Description

Described herein is a method for accommodating high priority calls on a congested communication link of a wide area or other communication network. In the following discussion, examples of specific embodiments of the present invention are set forth in order to provide the reader with a through understanding of the present invention. However, many of the details described below can be accomplished using equivalent methods or apparatus to those presented herein. Accordingly, the broader spirit and scope of the present invention is set forth in the claims that follow this detailed description and it is that broader spirit and scope (and not merely the specific examples set forth below) that should be recognized as defining the boundaries the present invention.

The basic mechanism adopted in the present invention is easily understood with reference to **Figure 2**. Process **200** describes a bandwidth renegotiation scheme for accommodating high priority calls over otherwise congested communication links in a wide area or other communications network. When a high priority call is received at a node of the network, a check is made to determine whether the call can be accommodated within the existing bandwidth utilization in the network.

For example, in the network **100** illustrated in **Figure 1**, if a high priority call is received at node **102**, controller **116** may be configured to determine whether the call can be accommodated on communication link **106** according to the existing bandwidth utilization thereof. If the call can be so accommodated, it is connected in the customary fashion. If the call cannot be so accommodated, however, the present bandwidth renegotiation scheme is employed to free up bandwidth on communication link **106** to the

point where the high priority call can be accommodated. Rather than simply dropping calls, by renegotiating the bandwidth utilization among existing calls, these calls as well as the new high priority call can be accommodated within network **100**.

To more fully appreciate the processes involved in the present scheme, it is helpful to understand how calls are handled in network **100** in accordance with the present invention. When a call is received from a CPE at node 102, it is mapped to an associated network address. For example, associated with controller 116 may be a database configured to provide appropriate mappings between dialed telephone numbers and network (e.g., ATM or Internet protocol (IP) addresses. An example of such a database **300** is shown in **Figure 3**. In this example, the telephone number (123) 456-7891 is mapped to the network address 123.221.456.78.

For the case of ATM network 100, this network address is associated with a virtual circuit (e.g., a permanent virtual circuit or PVC) between network nodes **102** and **104**, supported on communication link **106**. Thus, when the incoming call is parsed at node **102**, the dialed number may be extracted to determine the node for which the call is destined, according to the network address. In some cases, the dialed number and/or associated network address may be flagged as a high priority call. For example, the dialed number 911 may be flagged as a call of the highest possible priority (e.g., where more than two priority levels are available). In this way, incoming calls can be recognized as high priority or not.

As mentioned above, for voice calls a variety of compression schemes are available, depending upon the codec (coder-decoder) resources available at the end points of the communication link over which the call is transported. Each call (e.g., each

inbound TI channel at node 102) may negotiate for a particular compression scheme to be employed at the time a connection is established, or a default compression scheme may be used where no negotiation takes place. In accordance with the present invention, each voice port supported by the ATM interfaces **108** has a profile defining the available codec resources for that port. Incoming calls are mapped to these voice ports for communications across communication link **106** and in one embodiment, ATM interfaces **108** may each have up to 24 voice ports.

The profiles of the voice ports may be established by a network manager at the time PVCs are set up within network 100. Alternatively, or in addition, profiles may be exchanged between ports as part of a call set up process in the case of switched virtual circuits (SVCs). One mechanism for the exchange of such profiles is the ATM Adaptation Layer type 2 (AAL2) protocol. Recently, the ATM Forum has promulgated standards document af-vota-0113.000, entitled "ATM Trunk Networking Using AAL2 for Narrowband Services" (Feb. 1999). In that document, which is incorporated herein by reference as is set forth in its entirety, a scheme for selecting and managing encoding algorithms at nodes of an ATM network according to prearranged agreements is described. This scheme calls for the exchange of profile information in a manner suitable for use in accordance with the present invention.

By exchanging profiles, each node at an end of a communication link is aware of the codec (e.g., DSP) resources available at each corresponding voice port. Thus, the nodes can reach an agreement on which codec (i.e., which compression scheme) to use for a particular call. As noted above, compression schemes that provide high compression ratios tend to utilize less bandwidth than those that use low compression

ratios do. So, assuming that sufficient codec resources are available, when an incoming high priority call is recognized, if there is insufficient bandwidth available on communication link **106** to accommodate the new call, node **102** can instruct node 104 to adopt new compression schemes on one or more voice channels (i.e., calls) so as to free up bandwidth on communication link **106** to support the new call. This process can be repeated, as required, until no available codec resources remain.

The renegotiation process described above may be implemented using OAM (operations, administration and maintenance) cells that are exchanged between nodes 102 and 104. OAM cells are often exchanged between nodes of an ATM network and are used to convey a variety of information. In accordance with the present invention, the OAM cells are configured with payloads (e.g., cell type, function type and/or function specific fields) that contain instructions for moving to different compression schemes, and acknowledgements thereto. The format and use of OAM cells are well known in the art and need not be further described herein. What is unique is the use of such cells (which may be transmitted from a high priority queue within a node so as to ensure rapid call handling) to negotiate compression schemes for a channel on the fly.

Thus, a scheme for has been described. Although discussed with respect to specific embodiments, however, the broader applicability of the present invention should not be limited thereby. For example, although discussed with respect to the negotiation of compression schemes on the fly, other call parameters or connection parameters could be so negotiated on the fly. Thus, this broader applicability of the present invention is recited in the claims that follow.

Claims

What is claimed is:

1. A system, comprising a network node configured to negotiate for connections for high priority calls received at the node in the face of otherwise congested outbound communication links.
2. A system as in claim 1, wherein the negotiation is conducted in a fashion that will preserve connections for existing calls associated with the node.
3. A system as in claim 2, wherein the negotiation is conducted so as to cause one or more of the existing calls to consume less bandwidth over the outbound communication links than was consumed at a time prior to reception of the high priority calls.
4. A system as in claim 3, wherein the node is configured to initiate the negotiation depending on the availability of codec resources at the node.
5. A system as in claim 1, wherein the high priority calls comprise voice calls.
6. A system as in claim 1, wherein the node is configured to commence negotiations according to availability of codec resources at the node.
7. A system as in claim 1, wherein the node is configured to negotiate for one or more voice channels to accommodate the high priority calls depending upon selected compression schemes for existing calls transported on the outbound communication links.
8. A method comprising managing a communication link between nodes of a communication network so as to ensure connection availability for one or more high priority calls over the communication link through dynamic renegotiations of call parameters for existing calls transported over the communication link.
9. A method as in claim 8, wherein the calls are voice calls.

- 1 10. A method as in claim 9, wherein the communication link supports
2 communications according to the Asynchronous Transfer Mode.
- 1 11. A method as in claim 10, wherein the dynamic renegotiations comprise
2 negotiations of compression schemes for the voice calls.
- 1 12. A method as in claim 11, wherein the dynamic renegotiations are supported
2 according to codec availability at the nodes.
- 1 13. A method as in claim 12, wherein the dynamic renegotiations are accomplished
2 through the exchange of OAM cells between the nodes.
- 1 14. A method as in claim 12, wherein codec availability is determined according to
2 profile information maintained by the nodes.
- 1 15. A method as in claim 12, wherein the high priority calls are determined as such
2 according to database information regarding called numbers.
- 1 16. A network comprising:
2 a number of nodes connected through one or more communication links; and
3 a resource manager configured to allocate bandwidth over the communication
4 links to high priority calls received at one or more of the nodes without dropping existing
5 calls within the network.
- 1 17. The network of claim 16 wherein the resource manager allocates bandwidth
2 through dynamic renegotiations of existing bandwidth utilization within the network.
- 1 18. The network of claim 17 wherein the resource manager is a distributed resource
2 among the nodes of the network.
- 1 19. The network of claim 17 wherein the nodes each support multiple codec
2 resources, which compress voice information transmitted over the communication link.

- 1 20. The network of claim 17 wherein the dynamic renegotiations are supported
- 2 through the exchange of OAM cells between the nodes.

Abstract

A network node is configured to negotiate for connections for high priority calls (e.g., voice calls) received at the node in the face of otherwise congested outbound communication links. The negotiation is conducted in a fashion that will preserve connections for existing calls associated with the node. For example, the negotiation may be conducted so as to cause one or more of the existing calls to consume less bandwidth over the outbound communication links than was consumed at a time prior to reception of the high priority calls. Such negotiations may be initiated depending on the availability of codec resources and/or compression schemes at the node and may be conducted using OAM cells.



100

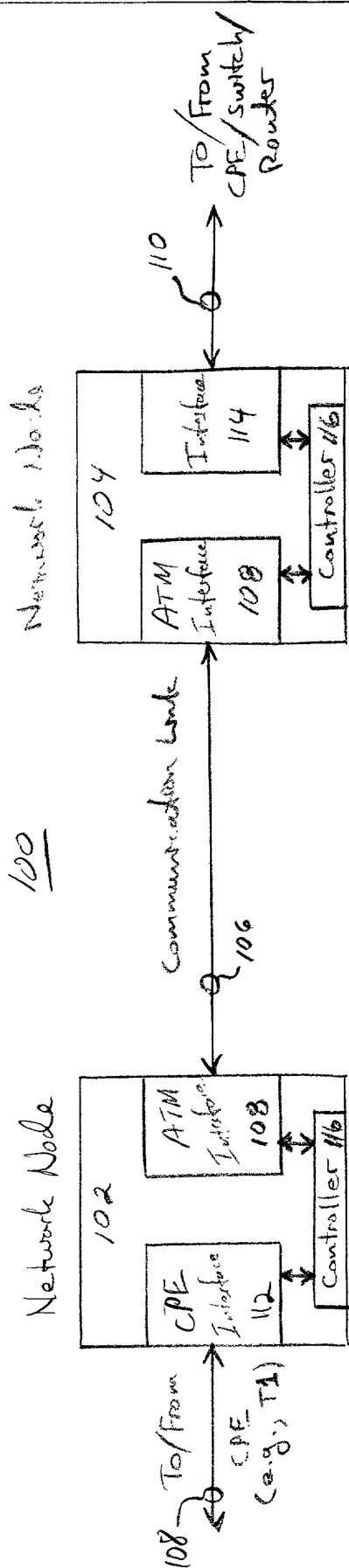


Fig. 1

Receive High Priority Call

200

Can the call be accommodated
within existing bandwidth utilization?

Yes

No

Accept call
& establish
connection in the
customary fashion.

Renegotiate bandwidth
utilization among
one or more existing
connections.

Fig. 2

dialed telephone number	network address	high Priority (Y/N)
(123) 456-7891 ⋮	123.221.456.78 ⋮	Y

300

Fig. 3

Attorney's Docket No.: 81862.P164

PATENT

DECLARATION AND POWER OF ATTORNEY FOR PATENT APPLICATION

As a below named inventor, I hereby declare that:

My residence, post office address and citizenship are as stated below, next to my name.

I believe I am the original, first, and sole inventor (if only one name is listed below) or an original, first, and joint inventor (if plural names are listed below) of the subject matter which is claimed and for which a patent is sought on the invention entitled

METHOD FOR SUPPORTING HIGH PRIORITY CALLS ON A CONGESTED WAN LINK

the specification of which

 X is attached hereto.
 was filed on _____ as
United States Application Number _____
or PCT International Application Number _____
and was amended on _____
(if applicable)

I hereby state that I have reviewed and understand the contents of the above-identified specification, including the claim(s), as amended by any amendment referred to above.

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<u>Prior Foreign Application(s)</u>			<u>Priority Claimed</u>	
<u>(Number)</u>	<u>(Country)</u>	<u>(Day/Month/Year Filed)</u>	<u>Yes</u>	<u>No</u>
<u>(Number)</u>	<u>(Country)</u>	<u>(Day/Month/Year Filed)</u>	<u>Yes</u>	<u>No</u>
<u>(Number)</u>	<u>(Country)</u>	<u>(Day/Month/Year Filed)</u>	<u>Yes</u>	<u>No</u>

I hereby claim the benefit under title 35, United States Code, Section 119(e) of any United States provisional application(s) listed below

(Application Number)	Filing Date
(Application Number)	Filing Date

I hereby claim the benefit under Title 35, United States Code, Section 120 of any United States application(s) listed below and, insofar as the subject matter of each of the claims of this application is not disclosed in the prior United States application in the manner provided by the first paragraph of Title 35, United States Code, Section 112, I acknowledge the duty to disclose all information known to me to be material to patentability as defined in Title 37, Code of Federal Regulations, Section 1.56 which became available between the filing date of the prior application and the national or PCT international filing date of this application:

(Application Number)	Filing Date	(Status -- patented, pending, abandoned)
(Application Number)	Filing Date	(Status -- patented, pending, abandoned)

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ZAFMAN LLP, 12400 Wilshire Boulevard 7th Floor, Los Angeles, California 90025 and direct
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(Name of Attorney or Agent)

I hereby declare that all statements made herein of my own knowledge are true and that all statements made on information and belief are believed to be true; and further that these statements were made with the knowledge that willful false statements and the like so made are punishable by fine or imprisonment, or both, under Section 1001 of Title 18 of the United States Code and that such willful false statements may jeopardize the validity of the application or any patent issued thereon.

Full Name of Sole/First Inventor Prasad Miriyala

Inventor's Signature *Prasad* Date 12 Oct 99

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(1) Prior art cited in search reports of a foreign patent office in a counterpart application, and

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(b) Under this section, information is material to patentability when it is not cumulative to information already of record or being made of record in the application, and

(1) It establishes, by itself or in combination with other information, a prima facie case of unpatentability of a claim; or

(2) It refutes, or is inconsistent with, a position the applicant takes in:

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(ii) Asserting an argument of patentability.

A prima facie case of unpatentability is established when the information compels a conclusion that a claim is unpatentable under the preponderance of evidence, burden-of-proof standard, giving each term in the claim its broadest reasonable construction consistent with the specification, and before any consideration is given to evidence which may be submitted in an attempt to establish a contrary conclusion of patentability.

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